Lecturer 6, 7, 8, 9 Radio ommunication Systems

- Introduction
- Types of Digital Modulation
  - Frequency Shift Keying FSK
    - MSK Minimum Shift Keying
  - Amplitude Shift Keying ASK
  - Phase Shift Keying PSK
- M-ary PSK Encoding
  - Quadrature QPSK
  - **8-PSK**
- QAM: (8-QAM)
- Carrier Recovery Circuits

# Digital Radio

#### Why Digital?

- Ease of processing,
- Ease of multiplexing, and
- Noise immunity.

#### All Digital Communications

Transmission, reception and processing of information.

#### Increasing of Information Capacity

No of independent symbols that can be carried through system in a given time.

## Information Capacity

- 1928 Hartley introduces useful rule:
  - Capacity C is proportional to both the bandwidth B and the time T:

$$C \sim BT$$

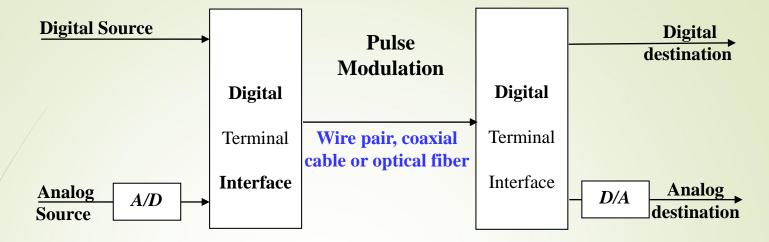
 $\rightarrow$  1948 Shannon published a limit for C:

$$C \leq B \log_2\left(1 + \frac{S}{N}\right) = 3.32 B \log_{10}\left(1 + \frac{S}{N}\right)$$

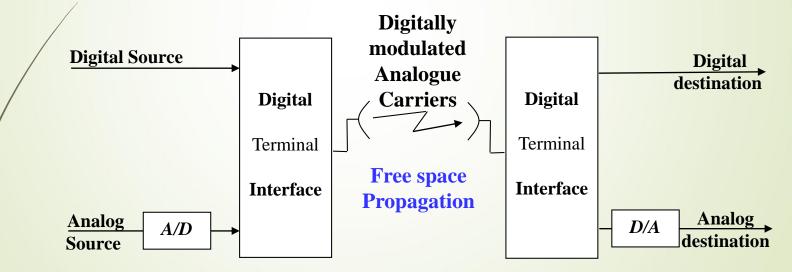
S/N = 1000 (30 dB), B = 2.7 kHz, C:  $C \le 2700 \log_2(1 + 1000) \le 26.9 kbps$ 

#### Limit Misunderstood

- Above example may be true, but it cannot be done with a binary system.
- To achieve 26.9 kbps through 2.7 kHz channel, each symbol must contain more than one bit of information.
- So, to achieve Shannon limit, digital transmission systems that have more than two output conditions (symbols) must be used.



#### (a) Baseband Transmission



(b) Digital Radio Transmission

# Types of Modulation

- Amplitude Shift Keying ASK تعديل إزاحة السعة
  - Frequency Shift Keying FSK تعديل إزاحة التردد -
    - تعديل الإزاحة الدنيا Minimum Shift Keying MSK تعديل الإزاحة الدنيا
- Gaussian Minimum Shift Keying GMSK تعديل الإزاحة الدنيا الجاوسي
  - Phase Shift Keying PSK تعديل إزاحة الطور
  - تعديل إزاحة الطور الثنائي Binary Phase Shift Keying BPSK.
  - تعديل إزاحة الطور التفاضلي Differential Phase Shift Keying DPSK.
  - راحة الطور متعدد المستويات M-ary Phase Shift Keying متعدد المستويات معديل إزاحة الطور متعدد المستويات
    - تعديل إزاحة الطور التعامدي Quadrature Phase Shift Keying QPSK.
      - تعديل إزاحة الطور الثماني Eight Levels Phase Shift Keying 8PSK تعديل إزاحة الطور الثماني
  - تعديل السعة التعامدي Quadrature Amplitude Modulation QAM

# FSK equency Shift Keying

# FSK Transmitter Signal Simple, low performance.

- Constant envelope angle modulation.

$$v_{FSK}(t) = V_c cos \left[ \left( \omega_c + f_m(t) \frac{\Delta \omega}{2} \right) t \right]$$

- $-f_m(t)$  binary digital modulating signal
- $V_c$ ,  $\omega_c$ , carrier amplitude, frequency
- $\blacksquare$  Carrier frequency shifts between  $\omega_c \pm \Delta\omega/2$
- $\blacksquare$  Shift rate equals the input bit rate  $f_b$  b/s.

## **FSK Transmitter**

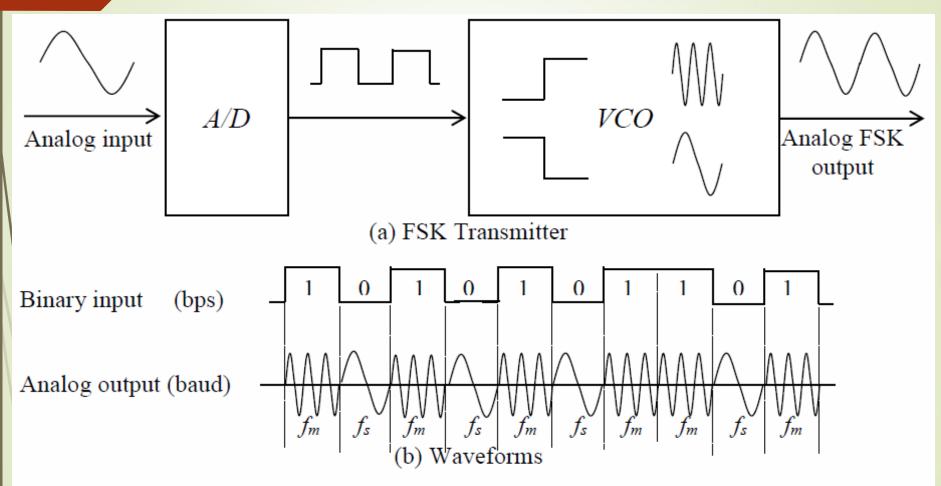
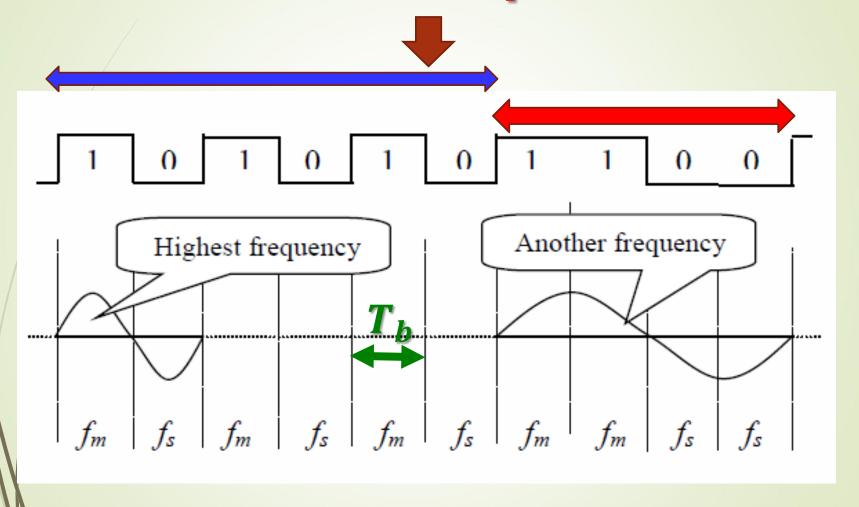


Fig 2.2 Ringry FSK Modulator

### Bit and Baud Rates

- Bit rate, in bits per second,
  - Is the rate of change at the input to the modulator.
- Baud rate, in symbol per sec
  - Is the rate of change at the output of the modulator and
  - Is equal to the reciprocal of the time of one output signaling element (termed as symbol).
- So, <u>baud</u> is the line speed in symbols per second.

### Possible frequencies



#### **13**

#### **Highest Modulating Frequency**

- If bit width is  $T_b$ , bit rate will be  $f_b = \frac{1}{T_b}$
- Fastest rate occurs when input is a series of alternating 1's and 0's:
- If fundamental frequency is considered, highest modulating frequency is one-half the input bit rate.

$$f_m = \frac{f_b}{2}$$

#### Modulation Index of FSK

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Peak frequency deviation  $\Delta f$  is one half the difference between  $f_m$  and  $f_s$ :

$$\Delta f = \frac{f_m - f_s}{2}$$

Formula for modulation index used in FM is also valid for binary FSK as:

$$MI = \frac{\Delta f}{f_m} = \frac{\frac{f_m - f_s}{2}}{\frac{f_b}{2}} = \frac{f_m - f_s}{f_b}$$

- MI is kept below 1.0 for narrow band FM.
- BW is determined from Bessel functions table.
- MI 0.5 and 1.0, either two or three sets of significant side frequencies are generated.
- Thus, minimum BW is two or three times the bit rate.

#### **Bandwidth of Binary FSK**

BW for FSK signal is given by Carson's rule in terms of the frequency deviation and the bandwidth of the digital modulation

$$BW_{FSK} = 2(\Delta f + B)$$

For alternating 1 and 0, the bandwidth equals the bit rate B = R:

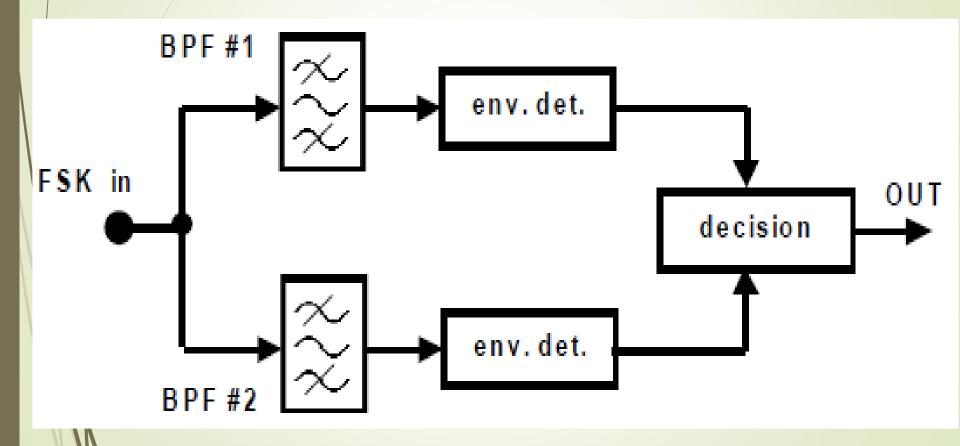
$$BW_{FSK} = 2(\Delta f + R)$$

# Receiver Binary FSK

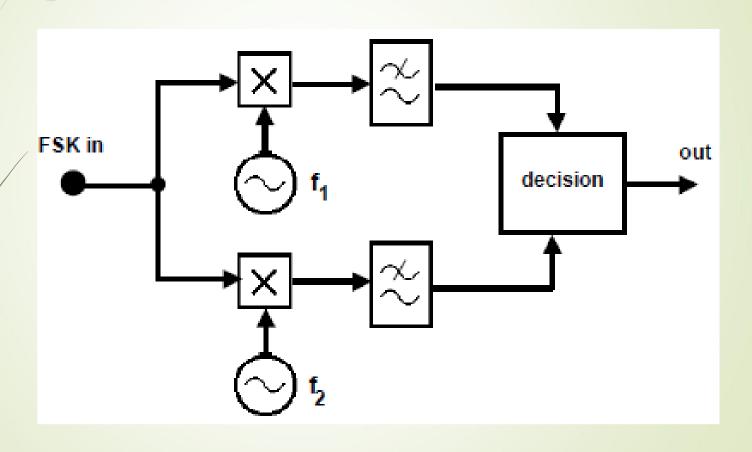
- Noncoherent Detection:
  - We do not have knowledge of the carrier.
  - Signal coming is divided into two BPF and envelope detectors.
  - Finally, binary restoration circuit.
- Coherent detection:
  - We need a complete knowledge of the exact carrier frequency on reception.
  - Received signal is applied into two multipliers, at f1 and f0, then to LPF.
  - Finally binary restoration circuit.

# Noncoherent Detector

**FSK** 



# FSK Synchronous Detector



# 19 Applications of FSK

- Binary FSK has a poorer error performance than PSK or QAM.
- Its use is restricted to lowperformance, low-cost, asynchronous data modems that are used for data communication over analogue, voice band telephone lines.

#### Bell 103-type FSK Modem

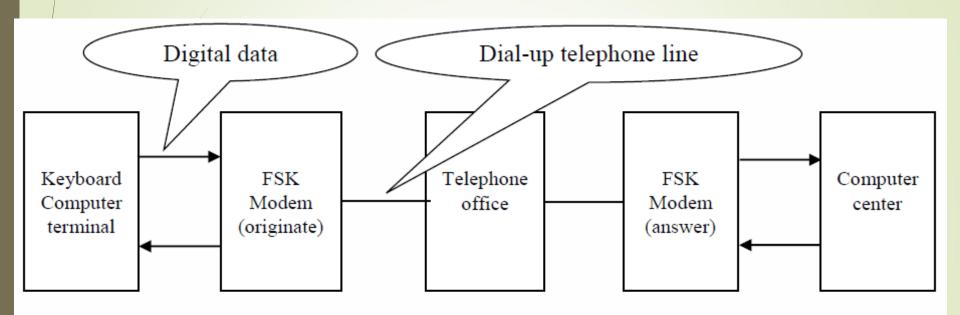


Table 2.1 Mark and Space Frequencies for the Bell Type 103 Modem

|             | Data             | Originate Modem         | Answer Modem            |
|-------------|------------------|-------------------------|-------------------------|
| Transmit    | Mark (binary 1)  | $f_1 = 1270 \text{ Hz}$ | $f_1 = 2225 \text{ Hz}$ |
| frequencies | Space (binary 0) | $f_2 = 1070 \text{ Hz}$ | $f_2 = 2025 \text{ Hz}$ |
| Receive     | Mark (binary 1)  | $f_1 = 2225 \text{ Hz}$ | $f_1 = 1270 \text{ Hz}$ |
| frequencies | Space (binary 0) | $f_2 = 2025 \text{ Hz}$ | $f_2 = 1070 \text{ Hz}$ |

#### Bell 103-type FSK Modem

- Keyboard-type computer terminals are often used for communication with a remote computer via dial-up telephone lines.
- Dial-up means that the computer terminal user calls the computer facility on a telephone and uses the telephone connection for data communication.
- Modem (modulator and demodulator) is connected to the phone line at each end as shown
- Two FSK frequency bands are used (one around 1 kHz and another around 2 kHz) so that it is possible to talk and listen simultaneously (full-duplex).
- The standard mark and space frequencies are shown in Table where the peak to peak deviation is  $2\Delta F = 200 \text{ Hz}$

# MSK Minimum Shift Keying

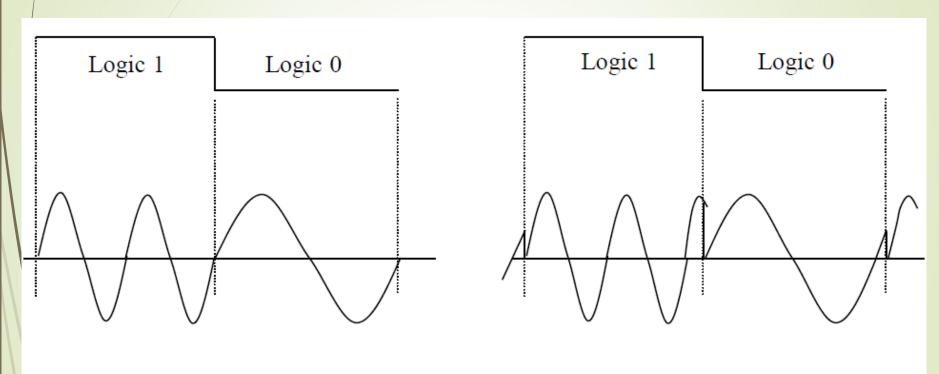
# Minimum Shift Keying

- ☐ MSK is a continuous phase FSK keying (CPFSK).
- MSK is FSK except mark and space frequencies are synchronized with input binary rate.
- Synchronous means precise time relationship.
- $\square$  Mark and space frequencies are separated from center frequency by odd multiple of one-half  $f_b$

$$f_m$$
 and  $f_s = n\frac{f_b}{2}$ 

- MSK has a better bit error performance than FSK for a given signal to noise ratio.
- MSK has less bandwidth than FSK
- □ However, it requires synchronizing circuits and is therefore more expensive to implement.

# MSK versus FSK



(a) Continuous Phase MSK

(b) Non-continuous FSK

Fig.2.8 Comparison of the Phase Continuity between MSK and FSK

# **ASK** Amplitude Shift Keying

### **Amplitude Shift Keying**

- ☐ In ASK, amplitude of carrier switches between; zero (Off state) and some amplitude (On state)
- ☐ Such systems are termed on-off-keyed systems OOK.
- Spectrum of OOK depend on the particular binary sequence to be transmitted. However, the amplitude modulated OOK is the DSB.SC given by:

$$f_{OOK}(t) = f_{ASK}(t) = A f(t) \cos \omega_c t$$

Spectrum of OOK signal is given as:

$$F_{OOK}(\omega) = F_{ASK}(\omega) = \frac{A}{2} [F(\omega - \omega_c) + F(\omega + \omega_c)]$$

# OOK Waveforms

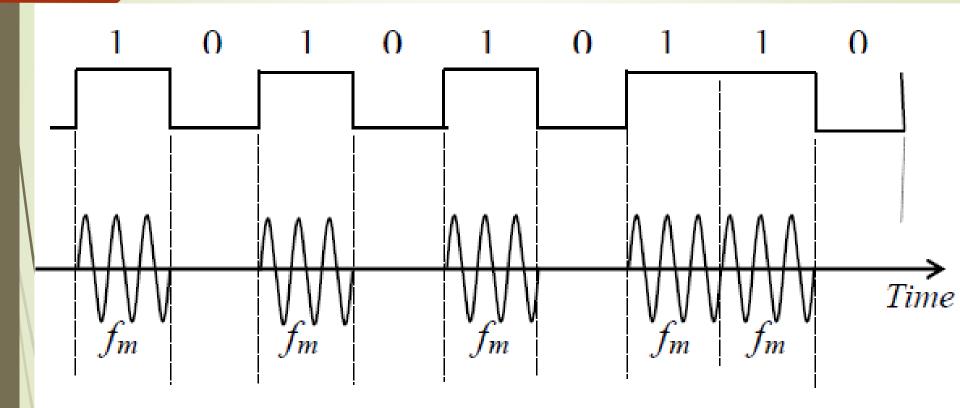


Fig.2.9 ASK or OOK Signal

### Spectrum of OOK

- Assume the digital signal f(t) is rectangular pulse (special case of binary in which all symbols are 0 except for one 1).
- For a pulse of amplitude A and duration 7, the spectrum of OOK modulator is given by:

$$F_{OOK}(\omega) = \frac{AT}{2} \left[ \frac{\sin(\omega - \omega_c)T/2}{(\omega - \omega_c)T/2} + \frac{\sin(\omega + \omega_c)T/2}{(\omega + \omega_c)T/2} \right] = \frac{AT}{2} \left[ Sa \left\{ \frac{(\omega - \omega_c)T}{2} \right\} + Sa \left\{ \frac{(\omega + \omega_c)T}{2} \right\} \right]$$

- For alternating 1's and 0's, spectrum is (sin x) /x.
- $\square$ So, spectrum of pulse of width T and period 2T which is translated to frequency  $f_c$  as in Fig.2.10

#### 29 Spectrum of Periodic OOK

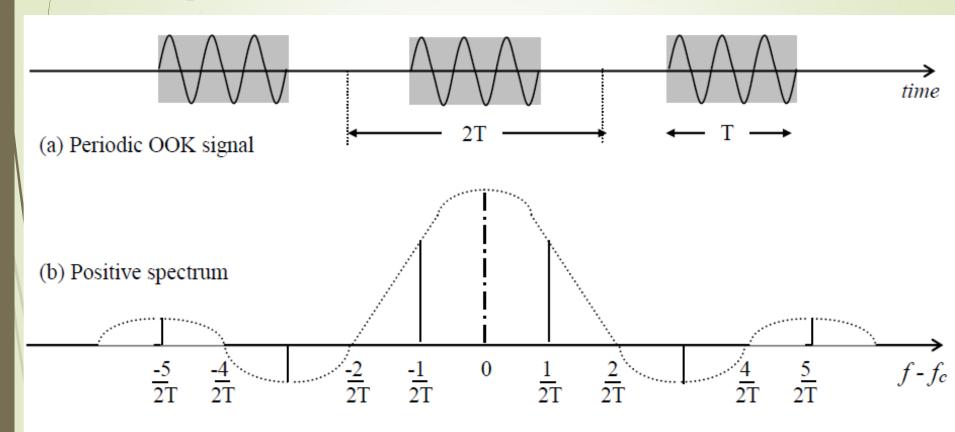


Fig.2.10 Spectrum of Periodic OOK Signal

# PSK Phase Shift Keying

# Phase Shift Keying

- □ PSK is similar to phase modulation PM except that its input gives rise to a limited number of output phases.
- With binary BPSK two output phases are possible for a single carrier frequency. One phase represents a løgic 1 and the other represents a logic 0.
- With carrier amplitude  $V_c$  and frequency  $\omega_c$  PSK voltage for binary digital modulating signal f(t) is:

$$v(t) = V_c f(t) \sin \omega_c t = \begin{cases} +V_c \sin \omega_c t & \text{if} & f(t) = +1 \\ -V_c \sin \omega_c t & \text{if} & f(t) = -1 \end{cases}$$

- So, the carrier amplitude remains constant, whereas its phase shifts by 180°.
- ☐ Recall, carrier phase shift rate equals input bit rate.

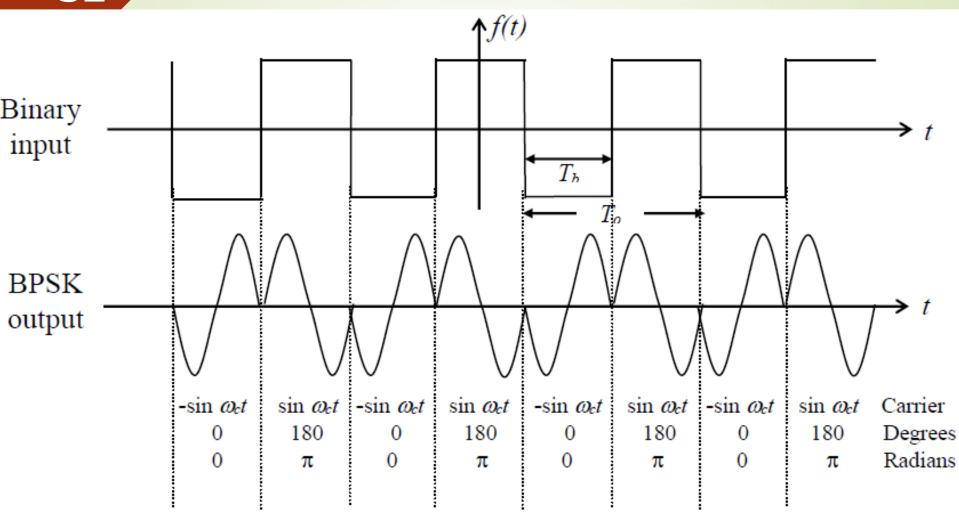
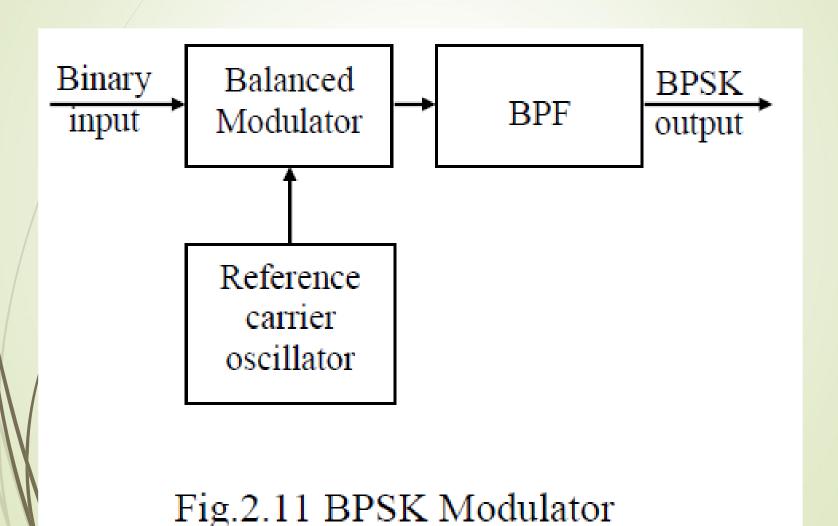


Fig.2.14 Output of BPSK Modulator

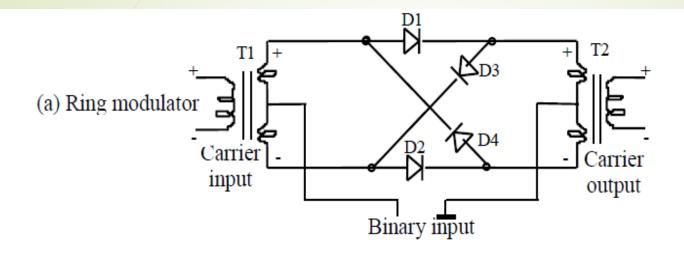
# **PSK Modulator**

- ☐ Simplified block diagram of BPSK is shown in Fig.2.11
- ☐ Balanced modulator acts as a phase reversing switch.
- Carrier is transferred to output either in phase or 180° with respect to reference carrier oscillator.
- ☐ Balanced ring modulator circuit is shown in Fig.2.12.
- Digital voltage must be much greater than the peak carrier voltage for proper operation.
  - □ For logic 1: D1 and D2 are ON while D3 and D4 are OFF, carrier voltage across T2 is in phase with the carrier voltage across T1 or the reference oscillator.
  - ☐ For logic 0: D1 and D2 are OFF while D3 and D4 are ON, carrier voltage across T2 is 180° out of phase with reference oscillator.

### Transmitter of BPSK



#### **PSK Balanced Ring Modulator**



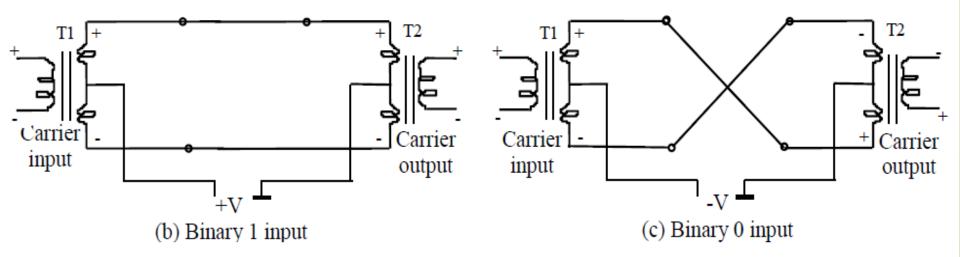


Fig.2.12 Balanced Ring Modulator

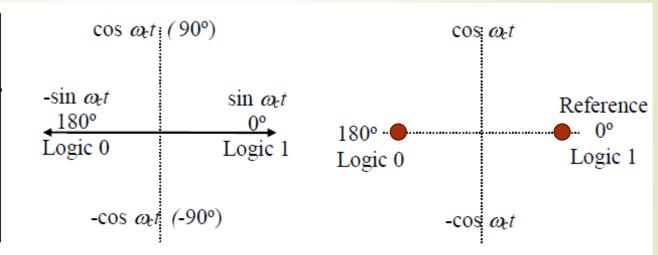
#### Representation of BPSK

- Figure shows truth table, phasor diagram and constellation diagram for a BPSK.
- Constellation diagram is similar to phasor except that the entire phasor is not drawn.
- Only the relative positions of the peaks of the phasors are shown.

#### **37**

#### Truth, Phasor, Constellation

| Binary<br>input | Output<br>phase |
|-----------------|-----------------|
| Logic 0         | 180°            |
| Logic 1         | 0°              |



(a) Truth table

(b) Phasor diagram

(c) Constellation diagram

Fig.2.13 BPSK Representation

#### **Band Width of PSK**

- 38
- □ Balanced modulator is product so carrier is multiplied by binary data (either +1 or −1).
- Also, widest bandwidth occurs when data is alternating 1/0 sequence.
- Product modulator output of the BPSK is:

$$output = \sin \omega_a t \sin \omega_c t = \frac{1}{2} \cos(\omega_c - \omega_a)t + \frac{1}{2} \cos(\omega_c + \omega_a)t$$

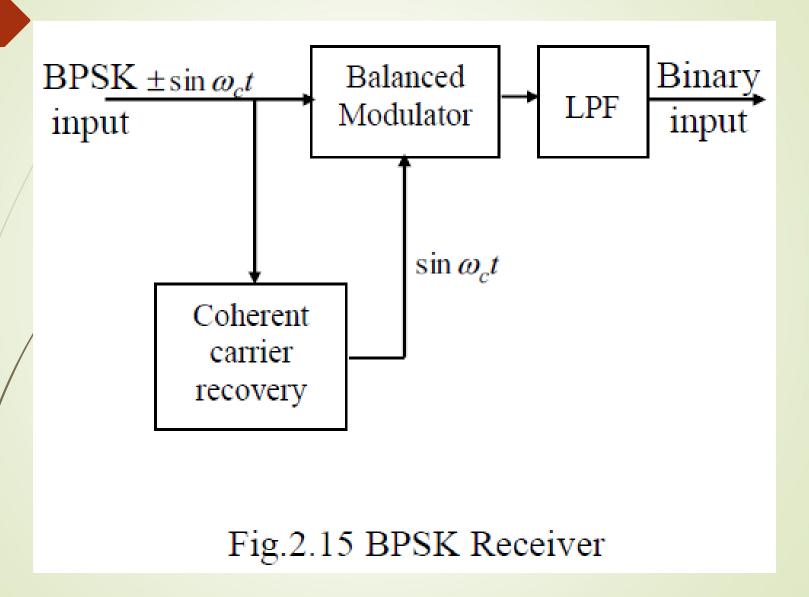
Consequently, minimum double-sided Nyquist bandwidth is:

$$B_{BPSK} = (\omega_c + \omega_a) - (\omega_c - \omega_a) = 2\omega_a = 2(f_b/2) = f_b$$

Minimum bandwidth to pass worst-case BPSK equals input bit rate. 39

- Simple block diagram of BPSK detection.
- Coherent carrier recovery circuit detects and regenerates carrier that is both frequency and phase coherent with the original transmit carrier.
- Balanced modulator output is the product of two inputs (BPSK signal and the recovered carrier).
- The LPF separates the recovered binary data from the complex demodulated spectrum.

#### **Detection of BPSK**



#### **Demodulation Process**

For input +sin w<sub>c</sub>t (logic 1), balanced output is:

$$Output = \sin \omega_c t \sin \omega_c t = \sin^2 \omega_c t = \frac{1}{2} (1 - \cos 2\omega_c t) = \frac{1}{2} - \frac{1}{2} \cos 2\omega_c t$$

$$Filter output = +\frac{1}{2} \quad dc \ voltage \equiv Logic1$$

• For input - $sin w_c t$  (logic 0), the output is:

$$Output = -\sin \omega_c t \sin \omega_c t = -\sin^2 \omega_c t = -\frac{1}{2} (1 - \cos 2\omega_c t) = -\frac{1}{2} + \frac{1}{2} \cos 2\omega_c t$$

$$Filter\ output = -\frac{1}{2} \quad dc\ voltage \equiv Logic\ 0$$

# M-ary Phase Shift Keying

# M-ary Encoding

- In M-ary, one of M possible signals may be transmitted during each signaling interval.
- It is advantageous to encode at a level higher than binary, e.g., 4PSK and 8PSK.
- Each possible transmitted signal of an M-ary message sequence is referred to as "symbol".
- Mathematically, the number of bits per symbol n is related to the number of possible signals M by:

$$M=2^n$$

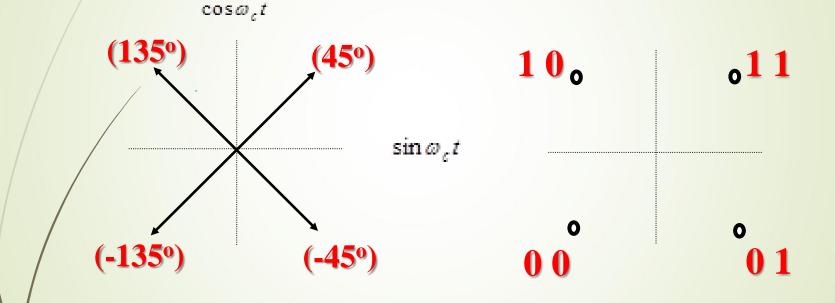
## Quadrature PSK

- QPSK, is another form of angle modulated, constant envelope digital modulation, and M=4 possible symbols.
- 4 phases are possible for a single carrier frequency.
- Binary input data are combined into groups of 2 bits called dibits.
- Éach dibit code generates one of the four possible output phases.
- For each 2- bit, a single output change occurs. So, the output baud rate is one-half of the input bit rate.

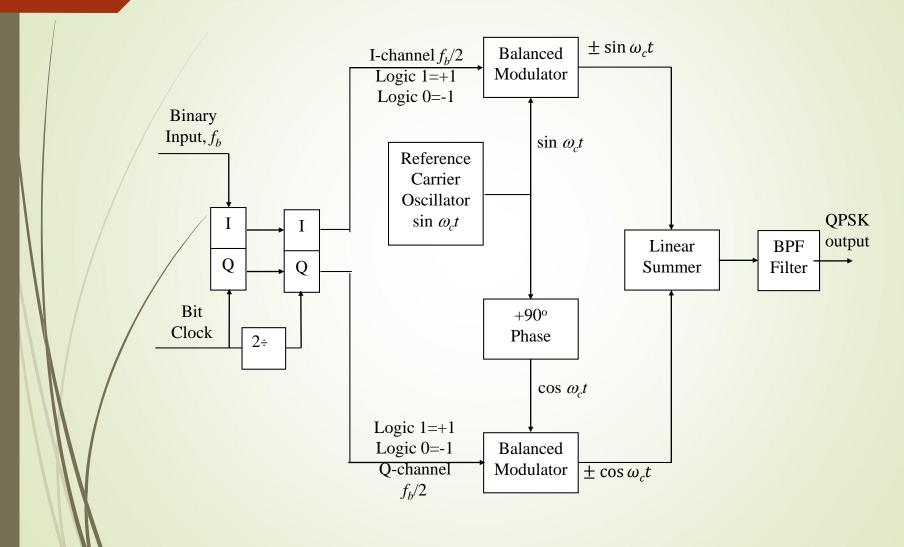
# QPSK Truth Table

| Inputs |   | Output |  |
|--------|---|--------|--|
| Α      | В | Phase  |  |
| 0      | 0 | -135   |  |
| 0      | 1 | -45    |  |
| 1      | 0 | +135   |  |
| 1      | 1 | +45    |  |

# QPSK Phasor Constellation



# **QPSK Transmitter**



### **Transmitter Operation**

- QPSK modulator is a two BPSK modulators combined in parallel.
- Two bits are clocked into the bit splitter.
- After both bits have been serially inputted, they are simultaneously parallel outputted.
- One bit is directed to I-channel to modulate the carrier that is in phase with the reference.
- Other bit is directed to Q-channel to modulate carrier that is 90° out of phase or in quadrature with the reference.
- If linear summer combines the two quadrature signals, there are 4 possible phases as follows:

 $\pm \sin \omega_c t \pm \cos \omega_c t$ 

#### Splitting to I and Q Channels

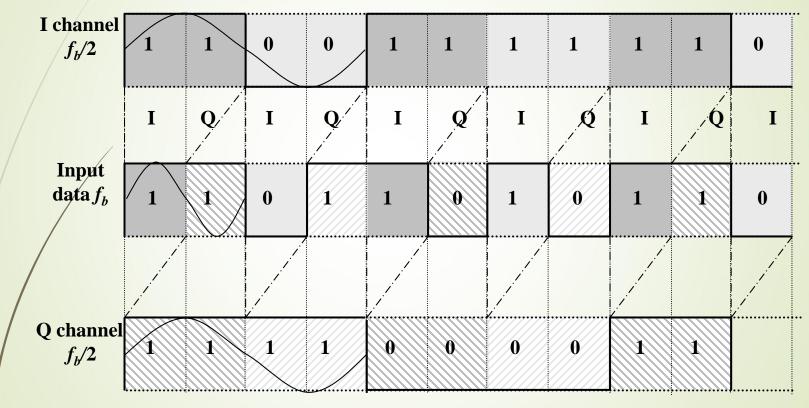


Fig.3.16: Highest Fundamental Frequency

#### **Bandwidth of QPSK**

#### **50**

- Input data rate  $f_b$  is divided into two channels.
- I or Q channel bit rate is  $\frac{1}{2}$  input rate, i.e.,  $f_b/2$ .
- Highest fundamental frequency at input of balanced modulators is one-fourth of input rate, i.e.,  $f_b/4$
- Balanced modulator product of I or Q channels:

**Output** = 
$$\sin \omega_a t \sin \omega_c t = \sin 2\pi \frac{f_b}{4} t \sin 2\pi f_c t$$

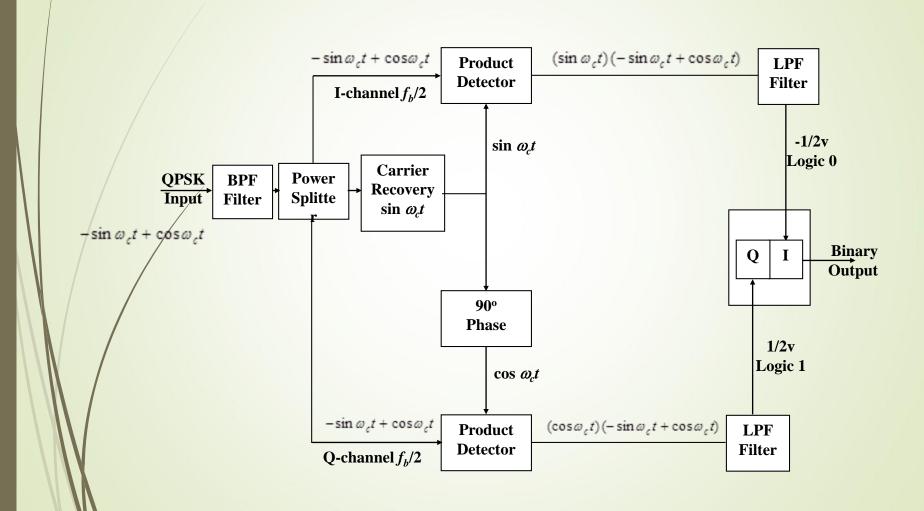
**Output** = 
$$\sin 2\pi \left( f_c - \frac{f_b}{4} \right) t \sin 2\pi \left( f_c + \frac{f_b}{4} \right) t$$

So, output extends from fc-fb/4 up to fc+fb/4:

$$BW_{QPSK} = f_c + \frac{f_b}{4} - \left(f_c - \frac{f_b}{4}\right) = \frac{f_b}{2}$$

Minimum bandwidth of QPSK is less than incoming rate so that bandwidth is compressed to  $f_b/2$  only.

# **QPSK Receiver**



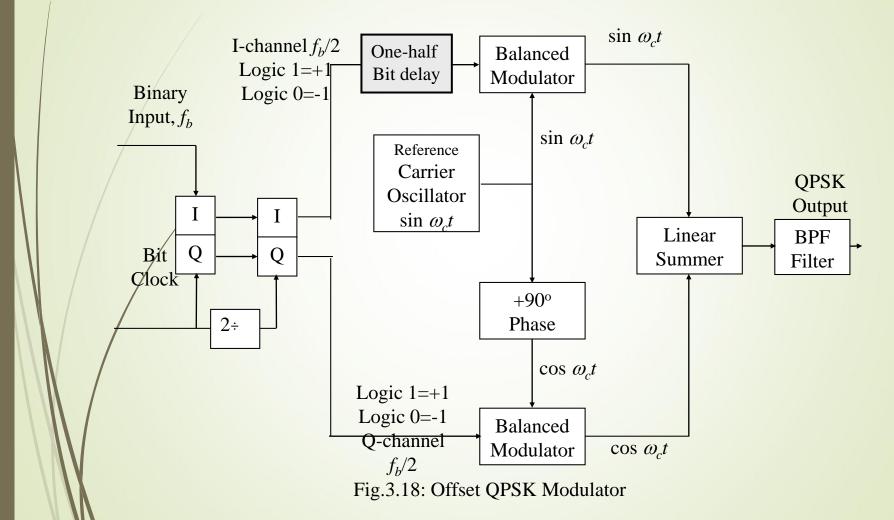
- Power splitter directs QPSK signal into I and Q channels and carrier recovery circuit.
- Carrier recovery circuit reproduces the original transmit reference carrier.
- QPSK signal is demodulated in I and Q channels through product detectors.
- Detectors outputs are fed to combining circuit, to convert from parallel I and Q channels to a single binary output.

#### Offset QPSK [OQPSK]

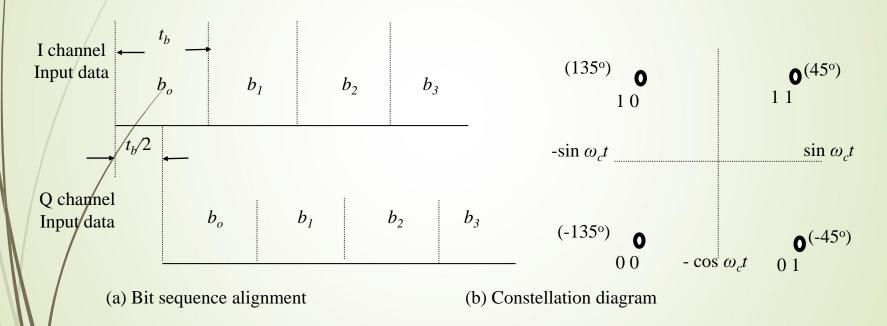
- A modified form of QPSK where the bit waveforms on I and Q channels are offset or shifted in phase by one-half of a bit time.
- It can be implemented by adding a delay.
- In QPSK, change from 00 to 11 or 01 to 10 causes 180° shift in output phase.
- Since changes in I channel of OQPSK occur at midpoints of Q bits, there is never more than a single bit change in the dibit code,
- So, 90° shift in phase improves performance.
- Disadvantage: changes in phase occur at twice the data rate so bandwidth is twice.

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### **OQPSK Transmitter**



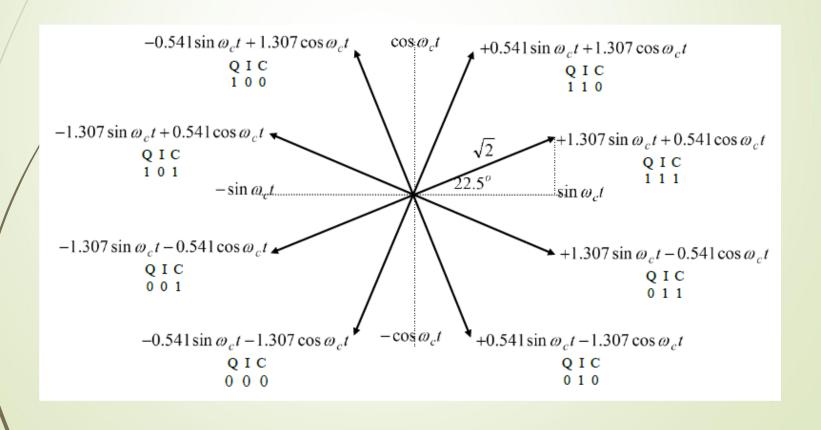
#### Offset Delay Concepts



**Fig.3.19: OQPSK** 

# 8 PSK

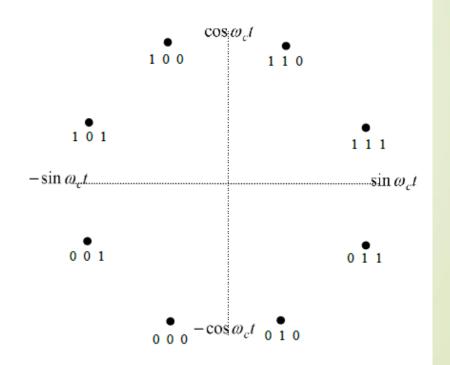
# 8PSK Phasor Diagram



# 8PSK Truth - Constellation

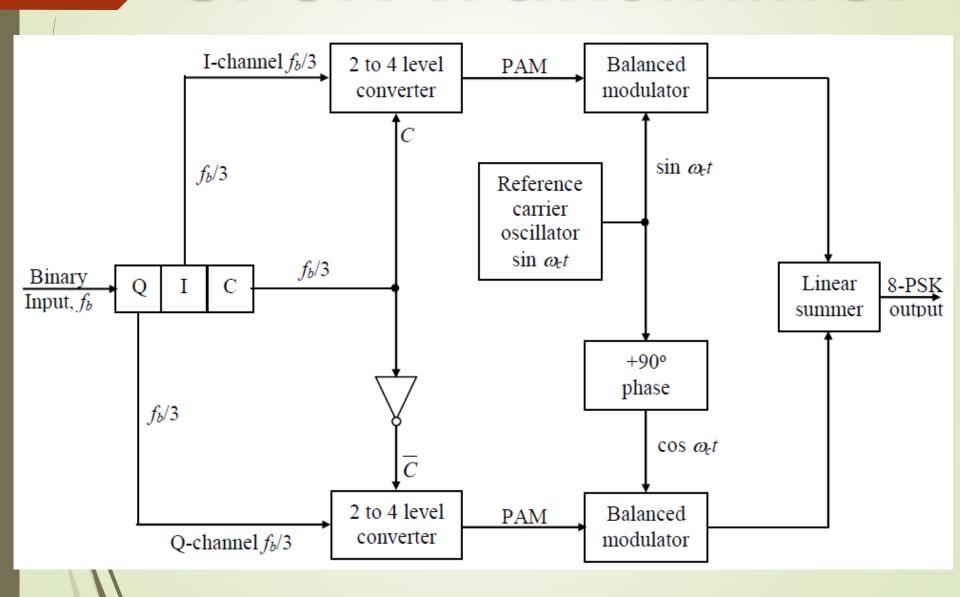
| Binary input |       |   | 8-PSK<br>Output |  |
|--------------|-------|---|-----------------|--|
| Q            | Q I C |   | phase           |  |
| 0            | 0     | 0 | -112.5          |  |
| 0            | 0     | 1 | -157.5          |  |
| 0            | 1     | 0 | -067.5          |  |
| 0            | 1     | 1 | -022.5          |  |
| 1            | 0     | 0 | +112.5          |  |
| 1            | 0     | 1 | +157.5          |  |
| 1            | 1     | 0 | +067.5          |  |
| 1            | 1     | 1 | +022.5          |  |

(b) The Truth Table of 8-PSK



(c) Constellation diagram of 8-PSK

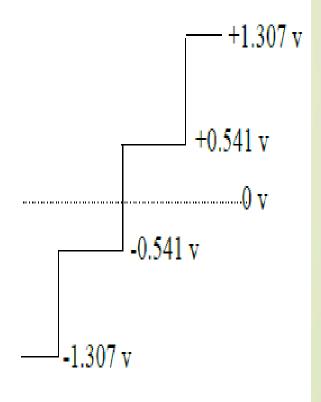
# **8PSK Transmitter**



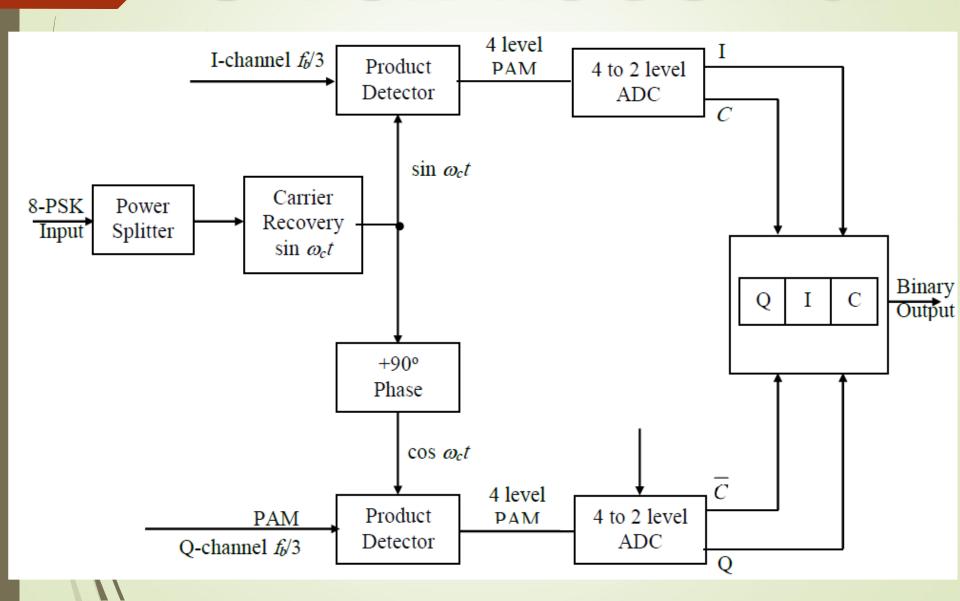
60

# 2-4 Levels Converter

| Ų | С           | Output                                   |
|---|-------------|--|
| 1 | l<br>0<br>1 | -1.307v<br>-0.541v<br>+1.307v<br>+0.541v |



# 8-PSK Receiver



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# QAM Quadrature Amplitude Modulation

# QAM

- QAM is a form of digital modulation, the information is contained in both the amplitude and the phase of the transmitted carrier.
- 8-QAM is M-ary encoding technique where M = 8.
- 8-QAM output is not a constantamplitude signal such as 8-PSK.

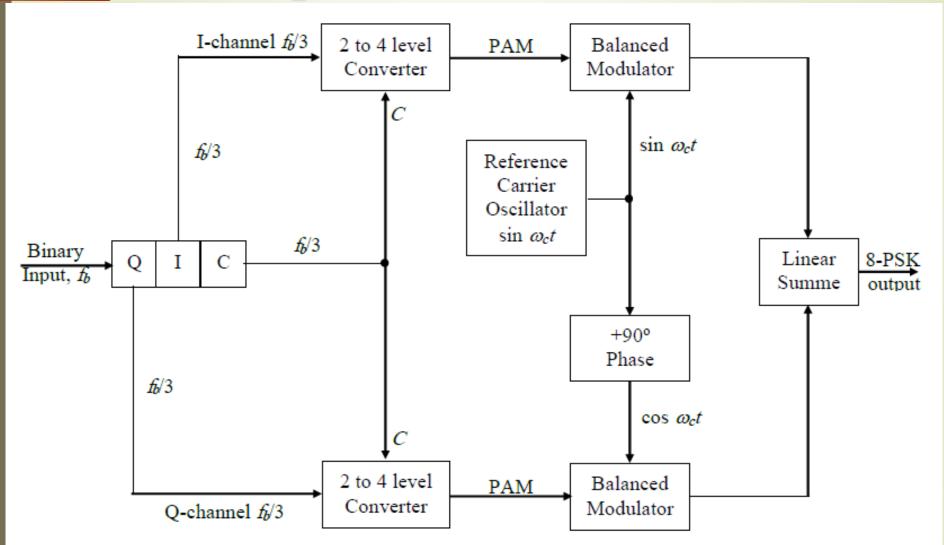
## 8-QAM Transmitter

- 8-QAM defers from 8-PSK in the inverter between the C and Q.
- Data are divided into I, Q, and C channels; each with a rate fb/3.
- I and Q bits determine the polarity of PAM signal at output of 2-to-4 level converters
- C channel determines the magnitude.
- 8-QAM output is not a constant-amplitude signal such as 8-PSK.

## 8-QAM Truth Table

| BINARY INPUT |   |   | 8-QAM OUTPUT |             |
|--------------|---|---|--------------|-------------|
| Q            | I | С | AMPLITUDE    | PHASE       |
| 0            | 0 | 0 | 0.765 V      | -135        |
| 0            | 0 | 1 | 1.848 V      | -135        |
| 0            | 1 | 0 | 0.765 V      | <b>-</b> 45 |
| 0            | 1 | 1 | 1.848 V      | <b>-</b> 45 |
| 1            | 0 | 0 | 0.765 V      | +135        |
| 1            | 0 | 1 | 1.848 V      | +135        |
| 1            | 1 | 0 | 0.765 V      | +45         |
| 1            | 1 | 1 | 1.848 V      | +45         |

## 8-QAM Transmitter



# Comparison

Table 2.2: Bandwidth Efficiency of Digital Modulation Techniques

| Modulation | Encoding   | Bandwidth, Hz     | Baud rate               | Efficiency, b/s/Hz |
|------------|------------|-------------------|-------------------------|--------------------|
| FSK        | Single bit | >f <sub>b</sub>   | $f_b$                   | <1                 |
| BPSK       | Single bit | $f_b$             | $f_b$                   | 1                  |
| QPSK       | Di-bit     | $f_b/2$           | $f_b/2$                 | 2                  |
| 8-PSK      | Tri-bit    | $f_b/3$           | $f_b/3$                 | 3                  |
| 8-QAM      | Tri-bit    | f <sub>b</sub> /3 | <i>f<sub>b</sub></i> /3 | 3                  |
| 16-PSK     | Quad-bit   | $f_b/4$           | $f_b/4$                 | 4                  |
| 16-QAM     | Quad-bit   | $f_b/4$           | $f_b/4$                 | 4                  |

# Squaring Loop

- The received BPSK signal is filtered to reduce the spectral width of noise.
- Squaring circuit removes the modulation and generates the second harmonic of carrier.
- This harmonic is phase tracked by PLL.
- The frequency of PLL (VCO output) is then divided by 2 and used as a phase reference for the product modulators.

#### **BPSK Carrier Recovery**

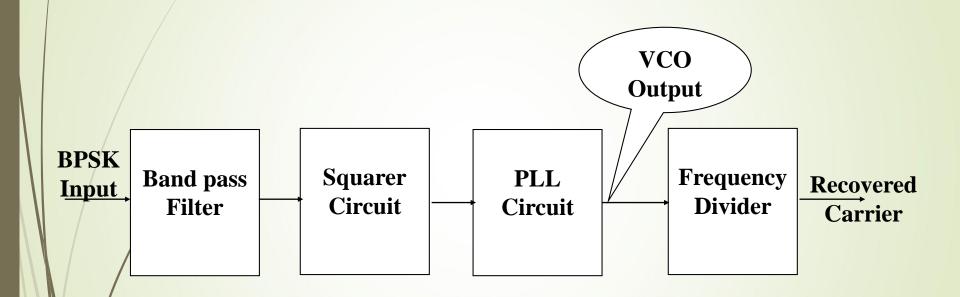


Fig.2.27 Squaring Loop Carrier Recovery for BPSK